

## Implementing IIR Digital Filters

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### INTRODUCTION

This application note describes the implementation of various digital filters using the PIC17C42, the first member of Microchip's 2nd generation of 8-bit microcontrollers. The PIC17C42 is a very high speed 8-bit microcontroller with an instruction cycle time of 250 ns (@ 16 MHz input clock). Even though the PIC17C42 is an 8-bit device, its high speed and efficient instruction set allows implementation of digital filters for practical applications. Traditionally digital filters have been implemented using expensive Digital Signal Processors (DSPs). In a system the DSP is normally a slave processor being controlled by either an 8-bit or 16-bit microcontroller. Where sampling rates are not high (especially in mechanical control systems), a single chip solution is possible using the PIC17C42.

This application note provides a few examples of implementing digital filters. Example code for 2nd order Infinite Impulse Response (IIR) filters is given. The following type of filters are implemented:

- Low Pass
- High Pass
- Band Pass
- Band Stop (notch) filter

This application note does not explain how to design a filter. Filter design theory is well established and is beyond the scope of this application note. It is assumed that a filter is designed according to the desired specifications. The desired digital filters may be designed using either standard techniques or using commonly available digital filter design software packages.

Finite Impulse Response (FIR) filters have many advantages over IIR filters, but are much more resource intensive (both in terms of execution time and RAM). On the other hand, IIR filters are quite attractive for implementing with the PIC17C42 resources. Especially where phase information is not so important, IIR filters are a good choice (FIR filters have a linear phase response). Of the various forms used for realizing digital filters (like, Direct form, Direct II form, Cascade form, Parallel, Lattice structure, etc.) the Direct II form is used in this application note. It is easy to understand and simple macros can be built using these structures.

### THEORY OF OPERATION

Digital filters in most cases assume the following form of relationship between the output and input sequences.

$$y(n) = \sum_{i=0}^M a_i y(n-i) + \sum_{j=0}^N b_j x(n-j)$$

The above equation basically states that the present output is a weighted sum of the past inputs and past outputs. In case of FIR filters, the weighted constants  $a_i = 0$  and in case of IIR filters, at least one of the  $a_i$  constants is non zero. In case of IIR, the above formula may be rewritten in terms of Z transform as:

$$H(z) = \frac{Y(z)}{X(z)} = \frac{\sum_{k=0}^M b_k z^{-k}}{1 + \sum_{k=1}^N a_k z^{-k}}$$

The above equation can further be rewritten in difference equation format as follows:

$$y(n) = \sum_{i=1}^M a_i y(n-i) + \sum_{j=0}^N b_j x(n-j)$$

Realization of the above equation is called the Direct Form II structure. For example, in case of a second order structure,  $M=N=2$ , gives the following difference equations:

## EQUATION 1:

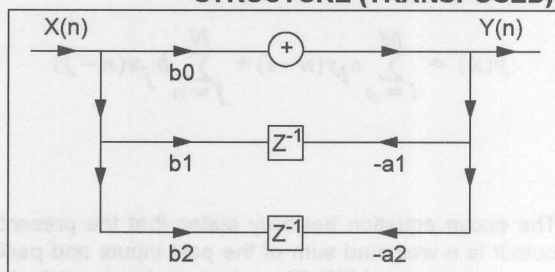
$$d(n) = x(n) + a_1 d(n-1) + a_2 d(n-2)$$

## EQUATION 2:

$$y(n) = b_0 d(n) + b_1 d(n-1) + b_2 d(n-2)$$

The above difference equations may be represented as shown in Figure 1.

**FIGURE 1: 2ND ORDER DIRECT FORM II STRUCTURE (TRANPOSED)**



The structure as shown in Figure 1 may be cascaded to attain a higher order filter. For example, if two stages are cascaded together, a 4th Order IIR Filter is obtained. This way, the output of the 1st stage becomes the input to the second stage. Multiple order filters are thus implemented by cascading a 2nd order filter structure as shown in Figure 1.

## IMPLEMENTATION

A 4th order IIR Filter is implemented by cascading two of the structures shown in Figure 1. The output  $Y$  (output of each filter stage) is computed by direct implementation of Equation 1 and Equation 2. Since each stage is similar algorithmically, it is implemented as a macro using Microchip's, Assembler/Linker for PIC17C42. This Macro (labelled BIQUAD) is called twice for implementing a 4th order filter. The output of the 1st stage is directly fed to the input of the second stage without any scaling.

Scaling may be required depending on the particular application. The user can modify the code very easily without any penalty on speed. Also, saturation arithmetic is not used. Overflows can be avoided by limiting the input sequence amplitude. All numbers are assumed to be 16 bits in Q15 format (15 decimal points, MSb is sign bit). Thus the user must scale and sign extend the input sequence accordingly. For example, if the input is from a 12-bit A/D converter, the user must sign extend the 12-bit input if bit 11 is a one.

The BIQUAD macro is a generic macro and can be used for all IIR filters whether it is Low Pass, High Pass, Band Pass or Band Stop. A general purpose 16x16

multiplier routine is also provided. This routine is implemented as a straight line code for speed considerations.

The 4th order IIR filter implemented is a Low Pass Filter with the specifications shown in Table 1.

**TABLE 1: FILTER CONSTANTS**

	BAND1	BAND2
Lower Band Edge	0.0	600 Hz
Upper Band Edge	500 Hz	1 kHz
Nominal Gain	1.0	0.0
Nominal Ripple	0.01	0.05
Maximum Ripple	0.00906	0.04601
Ripple in dB	0.07830	-26.75
Sampling Frequency = 2 kHz		

The Low Pass Filter is designed using a digital filter design package (DFDP™ by Atlanta Signal Processors Inc.). The filter package produces filter constants of the structure shown in Table 1. Table 2 shows the filter co-efficients that are obtained for the above Low Pass filter specification.

**TABLE 2: FILTER CO-EFFICIENTS**

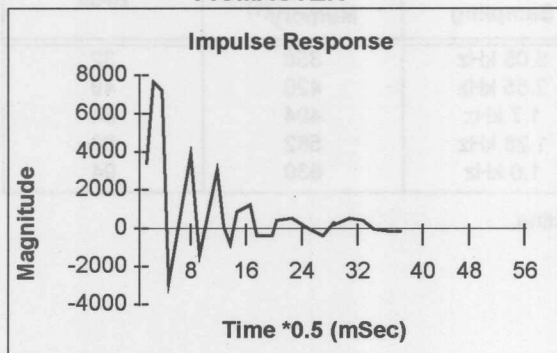
Stage	Co-efficients				
	a1	a2	b0	b1	b2
1	-0.133331	0.167145	0.285431	0.462921	0.285431
2	0.147827	0.765900	0.698273	0.499908	0.698273

The above filter co-efficients (5 per stage) are quantized to Q15 format (i.e they are multiplied by 32768) and saved in program memory (starting at label \_coeff\_lpass). The constants for both the stages are read into data memory using TLRD and TABLRD instructions in the Initialization Routine (labelled initFilter). The user may read the coefficients of only one stage at a time and save some RAM at the expense of speed.

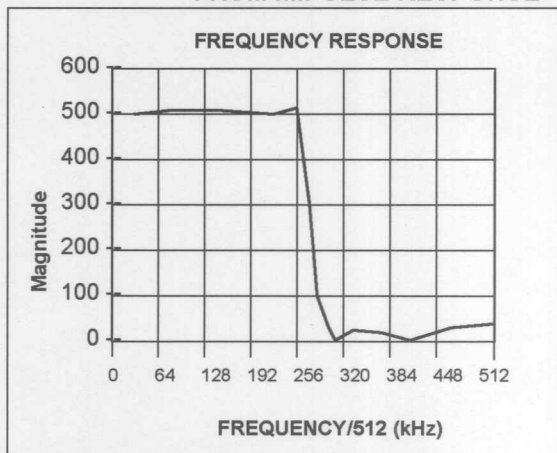
The sample 4th order Low Pass IIR Filter is tested by analyzing the impulse response of the filter. An impulse signal is fed as input to the filter. This is simulated by forcing the input to the filter by a large quantity (say 7F00h) on the first input sample, and the all zeros from the 2nd sample onwards. The output sequence is the filter's impulse response and is captured into the PICMASTER's (Microchip's Universal In-Circuit Emulator) real-time trace buffer. This captured data from PICMASTER is saved to file and analyzed. Analysis was done using MathCad™ for Windows® and a DSP Analysis program from Burr-Brown (DSPLAY™). The Fourier Transform of this impulse response of the filter should display the filter's frequency response, in this case being a Low Pass type. The plots of the impulse response and the frequency response are shown in Figure 2, Figure 3 and Figure 4.

MathCad is a trademark of MathSoft, Inc.  
DSPLAY is a trademark of Burr-Brown  
DFDP is a trademark of Atlanta Signal Processing Inc.  
Windows is a registered trademark of Microsoft Corporation.

**FIGURE 2: IMPULSE RESPONSE CAPTURED FROM PICMASTER**



**FIGURE 3: SPECTRUM COMPUTED FROM IMPULSE RESPONSE**



## PERFORMANCE

The resource requirements for filter implementations using a PIC17C42 is given in Table 3. These numbers can be used to determine whether a higher order filter can be executed in real-time. The same information may be used to determine the highest sampling rate possible.

## FILTER APPLICATIONS

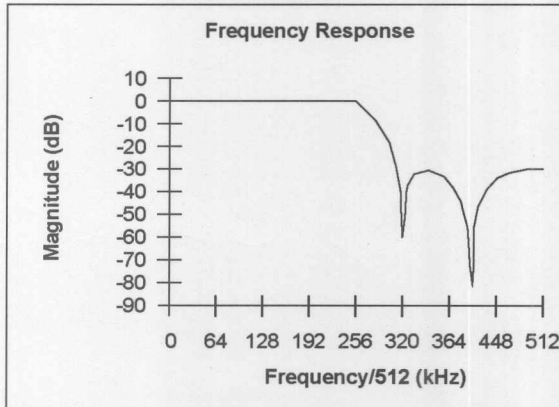
Digital filters find applications in many areas especially those involving processing of real world signals. In some applications like ABS systems in an automobile, digital filtering becomes a must. In this case elimination of noise (especially glitches and false readings of sensors) is very critical and thus becomes a requirement of digital signal processing.

**TABLE 3: RESOURCE REQUIREMENTS**

Timing (Cycles) <sup>(1)</sup>	#of Filter Stages*775 + 16
Program Memory (locations) <sup>(1)</sup>	#of Filter Stages*68 + 290
RAM (File Registers)	#of Filter Stages*16 + 16

Note 1: The above numbers do not include the initialization routine.

**FIGURE 4: LOG MAGNITUDE SPECTRUM OF IMPULSE RESPONSE**



Digital filters are also needed in Process Control where notch filters and low pass filters are desired because the signals from sensors are transmitted over long lines, especially in a very noisy environment. In these cases, typically a notch filter (centering 50 Hz or 60 Hz) is used. In cases of eliminating background noise, a band stop filter (e.g., 40 Hz to 120 Hz) is used. The sample code given in this application note can be used to design a feedback control system's digital compensator. For example, a typical DC Motor's digital compensator (like a dead-beat compensator) is of second order and has the same filter structure that is implemented in this application note.



# AN540

**TABLE 4: RESOURCE REQUIREMENTS**

Filter Order	Cycles	Real Time (@16 MHz)	Maximum Sampling	Program Memory <sup>(1)</sup>	RAM
2	791	197.75 $\mu$ s	5.05 kHz	358	32
4	1566	391.5 $\mu$ s	2.55 kHz	426	48
6	2341	585.25 $\mu$ s	1.7 kHz	494	64
8	3116	779.0 $\mu$ s	1.28 kHz	562	80
10	3891	972.75 $\mu$ s	1.0 kHz	630	94

Note 1: The above numbers do not include the initialization routine.

Please check the Microchip BBS for the latest version of the source code. Microchip's Worldwide Web Address: [www.microchip.com](http://www.microchip.com); Bulletin Board Support: MCHIPBBS using CompuServe® (CompuServe membership not required).

## APPENDIX A: IIR.LS0T

MPASM 01.40 Released

IIR.ASM 1-16-1997 14:48:37

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LOC  OBJECT CODE      LINE SOURCE TEXT
VALUE

00001          TITLE   "Digital IIR Filter Using PIC17C42"
00002
00003          LIST     P=17C42, columns=120, WRAP, R = DEC
00004
00005                      #include <pl7c42.inc>
00001          LIST
00002 ;Pl7C42.INC Standard Header File, Version 1.03 Microchip Technology, Inc.
00264          LIST
00006
00007          #define true      1
00008          #define false     0
00009          #define TRUE      1
00010          #define FALSE     0
00011
00012          #define LSB       0
00013          #define MSB       7
00014
00000001      00015  _INC      equ      1
00000000      00016  _NO_INC   equ      0
00000000      00017  _LOW      equ      0
00000001      00018  _HIGH     equ      1
00019
00020                      #include <17c42.mac>
00945          LIST
00021
00022                      #include <17c42iir.mac>
00001          LIST
00002 ;*****
00003 ;                      PIC17C42 MACRO
00004 ;
00005 ;Macro For A Bi-Quad IIR Filter
00006 ;      2nd order Direct Form (Transposed) Type
00007 ;
00008 ; Filter co-efficients B0 & B2 are assumed equal
00009 ;
00010 ; The difference equations for each cascade section is given by :
00011 ;      Y(n) = B0*D(n) + B1*D(n-1) + B2*D(n-2)
00012 ;      D(n) = X(n) - A1*D(n-1) - A2*D(n-2)
00013 ;      where X(n) = input sample, Y(n) = output of filter
00014 ;      and A1, A2, B0, B1, B2 are the Filter Co-efficients
00015 ;
00016 ; The above difference equations are only for 1 section of a
00017 ; 2nd order Direct_Form II Filter structure (IIR)
00018 ;
00019 ; NOTE :
00020 ;      It is possible to design the above structures
00021 ;      such that the co-efficients B0 = B2. If this is the
00022 ;      case,
00023 ;      Y(n) = B0*[D(n) + D(n-2)] = B2*[D(n) + D(n-2)]
00024 ;      This way, one multiplication can be avoided
00025 ;
00026 ; If a 4th order filter is to be implemented, the output of
00027 ; the 1st structure should be input to the 2nd cascade section
00028 ;

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00029 ;      Timing (WORST CASE) :
00030 ;      59+4*179 = 775 Cycles
00031 ;      (194 uS @ 16 Mhz)
00032 ;      Program Memory :
00033 ;      63 locations
00034 ;
00035 ;*****
00036 ;      The sample filters are designed so that B0=B2
00037 ;      This saves 1 multiplication
00038 ;
00000001 00039 B0_EQUALS_B2    equ      TRUE
00040 ;
00041 ;*****
00042 ;      Parameters to BIQUAD Macro :
00043 ;      Filter Constants A1, A2, B0, B1, B2
00044 ;      & D(n), D(n-1), D(n-2), filter stage #
00045 ;
00046 BIQUAD  MACRO    Ax1,Ax2,Bx0,Bx1,Dn,Dn_1,Dn_2,stage
00047 ;
00048 ;
00049 ; Compute Ax2*D(n-2)
00050 ;
00051 MOVFP16  Dn_2,AARG      ; D(n-2) = multiplier
00052 MOVFP16  Ax2,BARG       ; A2 = multiplicand
00053      call   DblMult      ; (ACCd,ACCc) = A2*D(n-2)
00054 ;
00055 ; Add product to output of 1st section
00056 ; Save result in 32 bit Accumulator
00057 ;
00058      ADD32    DPX,ACC
00059 ;
00060 ; Compute A1*D(n-1)
00061 ;
00062 MOVFP16  Dn_1,AARG      ; AARG = D(n-2) = multiplier
00063 MOVFP16  Ax1,BARG       ; BARG = A2 = multiplicand
00064      call   DblMult      ; (ACCd,ACCc) = A1*D(n-1)
00065 ;
00066 ; Compute A1*D(n-1) + A2*D(n-2) + output of previous section
00067 ;      multiplications already done, so simply perform a 32 bit add
00068 ;      of previously obtained multiplication results
00069 ;
00070      ADD32    DPX,ACC ; ACC = A1*D(n-1) + A2*D(n-2) + (output of 1st section)
00071 ;
00072 ;
00073 ; save the upper 16 bits of D(n) from the 32 bit accumulator
00074 ; left shift the result by 1, to adjust the decimal point after
00075 ; a Q15*Q15 multiplication
00076 ;
00077      rlcw     ACC+B1,w
00078      rlcw     ACC+B2,w
00079      movwf    Dn
00080      rlcw     ACC+B3,w      ; decimal adjust ( mult by 2)
00081      movwf    Dn+B1
00082 ;
00083 ; Compute B2 * [D(n) + D(n-2)]
00084 ;
00085      #if B0_EQUALS_B2
00086 ;
00087      ADD16ACC  Dn_2,Dn,AARG      ; AARG = Dn + D(n-2) = multiplier
00088 MOVFP16  Bx0,BARG      ; BARG = A2 = multiplicand
00089      call   DblMult      ; (ACCd,ACCc) = B2*[D(n)+D(n-2)]
00090 MOVFP32  DPX,ACC
00091 ;
00092      #else
00093 ;
00094 MOVFP16  Bx0,BARG

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00095 MOVFP16    Dn,AARG
00096          call    DblMult          ; B0*D(n)
00097 MOVFP32    DPX,ACC
00098
00099 MOVFP16    Bx2,BARG
00100 MOVFP16    Dn_2,AARG
00101          call    DblMult          ; B2*D(n-2)
00102          ADD32     DPX,ACC
00103
00104          #endif
00105 ;
00106 ; Shift down D(n-1) to D(n-2) after D(n-2) usage is no longer required.
00107 ; This way in the next iteration D(n-2) is equal to the present D(n-1)
00108 ;
00109          movfp     Dn_1,AARG+B0
00110          movpf     AARG+B0,Dn_2      ; Shift down D(n-1)
00111          movfp     Dn_1+B1,AARG+B1
00112          movpf     AARG+B1,Dn_2+B1  ; AARG = D(n-1) = multiplier
00113
00114 MOVFP16    Bx1,BARG          ; BARG = B1 = multiplicand
00115
00116          call    DblMult          ; (ACCd,ACCc) = B1*D(n-1)
00117 ;
00118 ; Compute Output Y = B1*D(n-1) + B2*D(n-2) + B0*D(n)
00119 ;                  = B1*D(n-1) + B0*[D(n) + D(n-2)]
00120 ; Since all multiplications are already done, simply perform a
00121 ; 32 bit addition
00122 ;
00123          ADD32     DPX,ACC ; ACC = B1*D(n-1) + B2*D(n-2) + B0*D(n)
00124 ;
00125 ; Shift down D(n) to D(n-1) so that in the next iteration, the new
00126 ; D(n-1) is the present D(n)
00127 ;
00128 MOV16      Dn,Dn_1 ; Shift down D(n) to D(n-1)
00129 ;
00130 ENDM
00131
00132 LIST
00023 ;
00024 ;*****
00025 ;          Second Order Direct Form IIR Filter
00026 ;
00027 ;
00028 ; In the code given below, a 4th order IIR Elliptic Lowpass Filter
00029 ; is implemented. Other order filters may be implemented by
00030 ; taking the following example code as a basis.
00031 ;
00032 ;
00033 ;          Program:          IIR.ASM
00034 ;          Revision Date:
00035 ;                          1-13-97          Compatibility with MPASWIN 1.40
00036 ;
00037 ;*****
00038 ;
00039 ;          The specifications of the filter are :
00040 ;
00041 ;          Sampling Frequency = 2.0 KHz
00042 ;
00043 ;          Filter Type = 4th Order Elliptic Lowpass Filter
00044 ;
00045 ;
00046 ;          Lower Band Edge  0.0          Band1          Band2
00047 ;          Upper Band Edge  500 Hz      600 Hz          1 KHz
00048 ;          Nominal Gain     1.0          0.0
00049 ;          Nominal Ripple   0.01         0.05
00050 ; Maximum Ripple  0.00906  0.04601

```



```

00051 ;      Ripple in dB      0.07830      -26.75
00052 ;
00053 ;      The Filter Co-efficients for the above specifications
00054 ; of the filter are computed as follows :
00055 ;
00056 ;      1st Section :
00057 ;                      A11 = -0.133331
00058 ;                      A12 = 0.167145
00059 ;                      B10 = 0.285431
00060 ;                      B11 = 0.462921
00061 ;                      B12 = 0.285431
00062 ;      2nd Section
00063 ;                      A21 = 0.147827
00064 ;                      A22 = 0.765900
00065 ;                      B20 = 0.698273
00066 ;                      B21 = 0.499908
00067 ;                      B22 = 0.698273
00068 ;
00069 ;
00070 ;      Performance (WORST Case):
00071 ;
00072 ;      Cycles      = #of Filter Stages*775 + 16
00073 ;                  = 2*775+16 = 1566 Cycles
00074 ;                  ( 391 uSec)
00075 ;      per each sample. Initialization
00076 ;      time after reset is not counted
00077 ;      Timing measured with B0_EQUALS_B2
00078 ;      set to TRUE (see BIQUAD Macro for
00079 ;      explanation)
00080 ;
00081 ;      Program Memory :
00082 ;                  = 16+ # of FilterStages * (BIQUAD Memory
00083 ;                  + filter co-efficients)
00084 ;                  + multiplier
00085 ;                  = 16+2*(63+5)+274 = 421 locations
00086 ;                  (excluding initialization)
00087 ;
00088 ;      RAM usage = 48 file registers
00089 ;      RAM usage/each additional stage = 16 file regs
00090 ;
00091 ;
00092 ;      This time is less than 2 Khz (500 uSec),
00093 ;      which means real time filtering is possible
00094 ;
00095 ; *****
00096 ; *****
00097 ; *****
00098 ;
00099 ;
00100 ;      CBLOCK 0
00101 ;      BB0, BB1, BB2, BB3
00102 ;      ENDC
00103 ; *****
00104 ;
00105 ;      CBLOCK 0x18
00106 ;      DPX, DPX1, DPX2, DPX3      ; arithmetic accumulator
00107 ;      AARG, AARG1, BARG, BARG1  ; multiply arguments
00108 ;      ENDC
00109 ;
00110 ;      CBLOCK
00111 ;      Dn1, Dn1_Hi
00112 ;      Dn1_1, Dn1_1_Hi
00113 ;      Dn1_2, Dn1_2_Hi
00114 ;
00115 ;      Dn2, Dn2_Hi
00116 ;      Dn2_1, Dn2_1_Hi

```



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0000002A 00117      Dn2_2, Dn2_2_Hi
00118      ENDC
00119
00120      CBLOCK
0000002C 00121      A11, A11_Hi
0000002E 00122      A12, A12_Hi
00000030 00123      B10, B10_Hi
00000032 00124      B11, B11_Hi      ; 1st Section Filter Co-efficients
00000034 00125      B12, B12_Hi
00126
00000036 00127      A21, A21_Hi
00000038 00128      A22, A22_Hi
0000003A 00129      B20, B20_Hi
0000003C 00130      B21, B21_Hi      ; 2nd Section Filter Co-efficients
0000003E 00131      B22, B22_Hi
00132      ENDC
00133
00134      CBLOCK
00000040 00135      X, X1      ; 16 bits of input stream
00000042 00136      Y, Y1      ; 16 bits of filter output
00137
00000044 00138      ACC, ACC1, ACC2, ACC3 ; 32 bit accumulator for computations
00139      ENDC
00140 ;
00000002 00141 FltStage .set 2
0000000A 00142 NumCoeff equ (5*FltStage) ; 5 Co-eff per stage
00143 ;
00000001 00144 F      equ 1
00145 ;
00000001 00146 LPASS .set TRUE
00000000 00147 HPASS .set FALSE
00000000 00148 BPASS .set FALSE      ; select the desired filter type
00000000 00149 BSTOP .set FALSE
00150 ;
00000001 00151 SIGNED equ TRUE      ; Set This To 'TRUE' for signed
00152 ;      ; multiplication and 'FALSE' for unsigned.
00153 ;
00154 ;*****
00155 ;      Test Program For Low Pass Filter
00156 ;*****
00157
0000      00158      ORG      0x0000
00159 ;
0000      00160 start
0000 E00D 00161      call      initFilter
0001 B000 00162      movlw     0x00
0002 0140 00163      movwf     X
0003 B07F 00164      movlw     0x7f      ; set initial Xn = X(0) = 0x7f00
0004 0141 00165      movwf     X+BB1      ; test for impulse response
00166 ;
0005      00167 NextPoint
0005 E022 00168      call      IIR_Filter
0006 A442 00169      tlwt      _LOW,Y
0007      00170 tracePoint
0007 AE43 00171      tablwt     _HIGH,0,Y+BB1
0008 0000 00172      nop
0009 2940 00173      clrf      X, F      ; set X(n) = 0 , n <> 0
000A 2941 00174      clrf      X+BB1, F      ; for simulating an Impulse
000B C005 00175      goto      NextPoint
00176 ;
000C C00C 00177 self      goto      self
00178 ;
00179 ;*****
00180 ;
000D      00181 initFilter
00182 ;

```

```

00183 ;   At first read the Filter Co-efficients from Prog. Mem to Data RAM
00184 ;
00185         #if      LPASS
000D B0C1 00186         movlw    LOW      _coeff_lpass
000E 010D 00187         movwf    TBLPTRL
000F B001 00188         movlw    HIGH     _coeff_lpass
0010 010E 00189         movwf    TBLPTRH
00190         #endif
00191         #if      HPASS
00192         movlw    LOW      _coeff_hpass
00193         movwf    TBLPTRL
00194         movlw    HIGH     _coeff_hpass
00195         movwf    TBLPTRH
00196         #endif
00197         #if      BPASS
00198         movlw    LOW      _coeff_bpass
00199         movwf    TBLPTRL
00200         movlw    HIGH     _coeff_bpass
00201         movwf    TBLPTRH
00202         #endif
00203         #if      BSTOP
00204         movlw    LOW      _coeff_bstop
00205         movwf    TBLPTRL
00206         movlw    HIGH     _coeff_bstop
00207         movwf    TBLPTRH
00208         #endif
00209
00210 ;
0011 B02C 00211         movlw    A11
0012 0101 00212         movwf    FSR0
0013 8404 00213         bsf     ALUSTA,FS0
0014 8D04 00214         bcf     ALUSTA,FS1           ; auto increment
00215 ;
00216 ; Read Filter Co-efficients from Program Memory
00217 ;
00218         movlw    NumCoeff
0015 B00A 00219         tablrd   _LOW,_INC,A11           ; garbage
0016 A92C 00220 NextCoeff
0017         tlrld    _LOW,INDF0
0017 A000 00221         tablrd   _HIGH,_INC,INDF0
0018 AB00 00222         decfsz   WREG, F
0019 170A 00223         goto     NextCoeff
001A C017 00224
00225 ;
00226 ; Initialize "Dn"s to zero
00227 ;
00228         movlw    Dn1
001B B020 00229         movwf    FSR0
001C 0101 00230         movlw    6*FltStage
001D B00C 00231 NextClr
001E         clrf    INDF0, F
001E 2900 00232         decfsz   WREG, F
001F 170A 00233         goto     NextClr
0020 C01E 00234
00235 ;
0021 0002 00236         return
00237 ;
00238 ;*****
00239 ;               1st Cascade Section
00240 ;*****
00241 ;
0022 00242 IIR_Filter
00243 ;
00244 ; Compute  $D(n) = X(n) + A1*D(n-1) + A2*D(n-2)$ 
00245 ; Since the filter constants are computed in Q15 format,
00246 ; X(n) must be multiplied by 2**15 and then added to the
00247 ; other terms.
00248 ;

```

```

00249 ; Move Input to accumulator after proper scaling
00250 ;
0022 8804 00251      bcf      ALUSTA,C
0023 1941 00252      rrcf     X+BB1, F
0024 1940 00253      rrcf     X, F
0025 290A 00254      clrf     WREG, F      ; Scale the input X
0026 190A 00255      rrcf     WREG, F
0027 0145 00256      movwf    ACC+BB1
0028 6A40 00257      movfp    X,WREG
0029 0146 00258      movwf    ACC+BB2
002A 6A41 00259      movfp    X+BB1,WREG
002B 0147 00260      movwf    ACC+BB3      ; ACC = scaled input : X*(2**15)
00261 ;
00262 ; 1st Biquad filter section
00263 ;
00264      BIQUAD   A11,A12,B10,B11,Dn1,Dn1_1,Dn1_2,1
M
M ;
M ; Compute Ax2*D(n-2)
M ;
M      MOVFP16   Dn1_2,AARG      ; D(n-2) = multiplier
M
002C 7C24 M      MOVFP    Dn1_2+B0,AARG+B0 ; move A(B0) to B(B0)
002D 7D25 M      MOVFP    Dn1_2+B1,AARG+B1 ; move A(B1) to B(B1)
M
M      MOVFP16   A12,BARG      ; A2 = multiplicand
M
002E 7E2E M      MOVFP    A12+B0,BARG+B0 ; move A(B0) to B(B0)
002F 7F2F M      MOVFP    A12+B1,BARG+B1 ; move A(B1) to B(B1)
M
0030 E0AF M      call     DblMult      ; (ACCd,ACCc) = A2*D(n-2)
M ;
M ; Add product to output of 1st section
M ; Save result in 32 bit Accumulator
M ;
M      ADD32     DPX,ACC
M
0031 6A18 M      MOVFP    DPX+B0,WREG      ; get lowest byte of a into w
0032 0F44 M      ADDWF     ACC+B0, F      ; add lowest byte of b, save in b(B0)
0033 6A19 M      MOVFP    DPX+B1,WREG      ; get 2nd byte of a into w
0034 1145 M      ADDWFC    ACC+B1, F      ; add 2nd byte of b, save in b(B1)
0035 6A1A M      MOVFP    DPX+B2,WREG      ; get 3rd byte of a into w
0036 1146 M      ADDWFC    ACC+B2, F      ; add 3rd byte of b, save in b(B2)
0037 6A1B M      MOVFP    DPX+B3,WREG      ; get 4th byte of a into w
0038 1147 M      ADDWFC    ACC+B3, F      ; add 4th byte of b, save in b(B3)
M
M ;
M ; Compute A1*D(n-1)
M ;
M      MOVFP16   Dn1_1,AARG      ; AARG = D(n-2) = multiplier
M
0039 7C22 M      MOVFP    Dn1_1+B0,AARG+B0 ; move A(B0) to B(B0)
003A 7D23 M      MOVFP    Dn1_1+B1,AARG+B1 ; move A(B1) to B(B1)
M
M      MOVFP16   A11,BARG      ; BARG = A2 = multiplicand
M
003B 7E2C M      MOVFP    A11+B0,BARG+B0 ; move A(B0) to B(B0)
003C 7F2D M      MOVFP    A11+B1,BARG+B1 ; move A(B1) to B(B1)
M
003D E0AF M      call     DblMult      ; (ACCd,ACCc) = A1*D(n-1)
M ;
M ; Compute A1*D(n-1) + A2*D(n-2) + output of previous section
M ; multiplications already done, so simply perform a 32 bit add
M ; of previously obtained multiplication results
M ;
M      ADD32     DPX,ACC      ; ACC = A1*D(n-1) + A2*D(n-2) +

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M                                     ; (output of 1st section)
003E 6A18 M MOVFP DPX+B0,WREG ; get lowest byte of a into w
003F 0F44 M ADDWF ACC+B0, F ; add lowest byte of b, save in b(B0)
0040 6A19 M MOVFP DPX+B1,WREG ; get 2nd byte of a into w
0041 1145 M ADDWFC ACC+B1, F ; add 2nd byte of b, save in b(B1)
0042 6A1A M MOVFP DPX+B2,WREG ; get 3rd byte of a into w
0043 1146 M ADDWFC ACC+B2, F ; add 3rd byte of b, save in b(B2)
0044 6A1B M MOVFP DPX+B3,WREG ; get 4th byte of a into w
0045 1147 M ADDWFC ACC+B3, F ; add 4th byte of b, save in b(B3)
M
M ;
M ;
M ; save the upper 16 bits of D(n) from the 32 bit accumulator
M ; left shift the result by 1, to adjust the decimal point after
M ; a Q15*Q15 multiplication
M ;
0046 1A45 M rlcfc ACC+B1,w
0047 1A46 M rlcfc ACC+B2,w
0048 0120 M movwf Dn1
0049 1A47 M rlcfc ACC+B3,w ; decimal adjust ( mult by 2)
004A 0121 M movwf Dn1+B1
M ;
M ; Compute B2 * [D(n) + D(n-2)]
M ;
M #if B0_EQUALS_B2
M
M ADD16ACC Dn1_2,Dn1,AARG ; AARG = Dn + D(n-2) = multiplier
M
004B 6A24 M movfp Dn1_2+B0,WREG
004C 0E20 M addwf Dn1+B0,w
004D 011C M movwf AARG+B0
004E 6A25 M movfp Dn1_2+B1,WREG
004F 1021 M addwfc Dn1+B1,w
0050 011D M movwf AARG+B1
M
M MOVFP16 B10,BARG ; BARG = A2 = multiplicand
M
0051 7E30 M MOVFP B10+B0,BARG+B0 ; move A(B0) to B(B0)
0052 7F31 M MOVFP B10+B1,BARG+B1 ; move A(B1) to B(B1)
M
0053 E0AF M call DblMult ; (ACCd,ACCc) = B2*[D(n)+D(n-2)]
M MOVFP32 DPX,ACC
M
0054 5844 M MOVFP DPX+B0,ACC+B0 ; move A(B0) to B(B0)
0055 5945 M MOVFP DPX+B1,ACC+B1 ; move A(B1) to B(B1)
0056 5A46 M MOVFP DPX+B2,ACC+B2 ; move A(B2) to B(B2)
0057 5B47 M MOVFP DPX+B3,ACC+B3 ; move A(B3) to B(B3)
M
M
M #else
M
M MOVFP16 B10,BARG
M MOVFP16 Dn1,AARG
M call DblMult ; B0*D(n)
M MOVFP32 DPX,ACC
M
M MOVFP16 Bx2,BARG
M MOVFP16 Dn1_2,AARG
M call DblMult ; B2*D(n-2)
M ADD32 DPX,ACC
M
M #endif
M ;
M ; Shift down D(n-1) to D(n-2) after D(n-2) usage is no longer required.
M ; This way in the next iteration D(n-2) is equal to the present D(n-1)
M ;

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0058 7C22      M    movfp    Dn1_1,AARG+B0
0059 5C24      M    movpf    AARG+B0,Dn1_2      ; Shift down D(n-1)
005A 7D23      M    movfp    Dn1_1+B1,AARG+B1
005B 5D25      M    movpf    AARG+B1,Dn1_2+B1      ; AARG = D(n-1) = multiplier
M
M    MOVFP16    B11,BARG      ; BARG = B1 = multiplicand
M
005C 7E32      M    MOVFP    B11+B0,BARG+B0      ; move A(B0) to B(B0)
005D 7F33      M    MOVFP    B11+B1,BARG+B1      ; move A(B1) to B(B1)
M
M
005E E0AF      M    call     DblMult      ; (ACCd,ACCc) = B1*D(n-1)
M ;
M ; Compute Output Y = B1*D(n-1) + B2*D(n-2) + B0*D(n)
M ;               = B1*D(n-1) + B0*[D(n) + D(n-2)]
M ; Since all multiplications are already done, simply perform a
M ; 32 bit addition
M ;
M    ADD32     DPX,ACC ; ACC = B1*D(n-1) + B2*D(n-2) + B0*D(n)
M
005F 6A18      M    MOVFP    DPX+B0,WREG      ; get lowest byte of a into w
0060 0F44      M    ADDWF    ACC+B0, F      ; add lowest byte of b, save in b(B0)
0061 6A19      M    MOVFP    DPX+B1,WR      ; get 2nd byte of a into w
0062 1145      M    ADDWFC    ACC+B1, F      ; add 2nd byte of b, save in b(B1)
0063 6A1A      M    MOVFP    DPX+B2,WREG      ; get 3rd byte of a into w
0064 1146      M    ADDWFC    ACC+B2, F      ; add 3rd byte of b, save in b(B2)
0065 6A1B      M    MOVFP    DPX+B3,WREG      ; get 4th byte of a into w
0066 1147      M    ADDWFC    ACC+B3, F      ; add 4th byte of b, save in b(B3)
M
M ;
M ; Shift down D(n) to D(n-1) so that in the next iteration, the new
M ; D(n-1) is the present D(n)
M ;
M    MOV16     Dn1,Dn1_1 ; Shift down D(n) to D(n-1)
M
0067 6A20      M    MOVFP    Dn1+B0,WREG      ; get byte of a into w
0068 0122      M    MOVWF    Dn1_1+B0      ; move to b(B0)
0069 6A21      M    MOVFP    Dn1+B1,WREG      ; get byte of a into w
006A 0123      M    MOVWF    Dn1_1+B1      ; move to b(B1)
M
M ;
00265 ;
00266 ; 2nd Biquad filter section
00267 ;
00268 BIQUAD    A21,A22,B20,B21,Dn2,Dn2_1,Dn2_2,2
M
M ;
M ; Compute Ax2*D(n-2)
M ;
M    MOVFP16    Dn2_2,AARG      ; D(n-2) = multiplier
M
006B 7C2A      M    MOVFP    Dn2_2+B0,AARG+B0      ; move A(B0) to B(B0)
006C 7D2B      M    MOVFP    Dn2_2+B1,AARG+B1      ; move A(B1) to B(B1)
M
M    MOVFP16    A22,BARG      ; A2 = multiplicand
M
006D 7E38      M    MOVFP    A22+B0,BARG+B0      ; move A(B0) to B(B0)
006E 7F39      M    MOVFP    A22+B1,BARG+B1      ; move A(B1) to B(B1)
M
M
006F E0AF      M    call     DblMult      ; (ACCd,ACCc) = A2*D(n-2)
M ;
M ; Add product to output of 1st section
M ; Save result in 32 bit Accumulator
M ;
M    ADD32     DPX,ACC
M

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0070 6A18      M    MOVFP    DPX+B0,WREG          ; get lowest byte of a into w
0071 0F44      M    ADDWF    ACC+B0, F          ; add lowest byte of b, save in b(B0)
0072 6A19      M    MOVFP    DPX+B1,WREG          ; get 2nd byte of a into w
0073 1145      M    ADDWFC    ACC+B1, F          ; add 2nd byte of b, save in b(B1)
0074 6A1A      M    MOVFP    DPX+B2,WREG          ; get 3rd byte of a into w
0075 1146      M    ADDWFC    ACC+B2, F          ; add 3rd byte of b, save in b(B2)
0076 6A1B      M    MOVFP    DPX+B3,WREG          ; get 4th byte of a into w
0077 1147      M    ADDWFC    ACC+B3, F          ; add 4th byte of b, save in b(B3)
M
M ;
M ; Compute A1*D(n-1)
M ;
M    MOVFP16   Dn2_1,AARG          ; AARG = D(n-2) = multiplier
M
0078 7C28      M    MOVFP    Dn2_1+B0,AARG+B0      ; move A(B0) to B(B0)
0079 7D29      M    MOVFP    Dn2_1+B1,AARG+B1      ; move A(B1) to B(B1)
M
M    MOVFP16   A21,BARG          ; BARG = A2 = multiplicand
M
007A 7E36      M    MOVFP    A21+B0,BARG+B0      ; move A(B0) to B(B0)
007B 7F37      M    MOVFP    A21+B1,BARG+B1      ; move A(B1) to B(B1)
M
007C E0AF      M    call     DblMult          ; (ACCd,ACCc) = A1*D(n-1)
M ;
M ; Compute A1*D(n-1) + A2*D(n-2) + output of previous section
M ; multiplications already done, so simply perform a 32 bit add
M ; of previously obtained multiplication results
M ;
M    ADD32     DPX,ACC          ; ACC = A1*D(n-1) + A2*D(n-2) +
M                                ; (output of 1st section)
007D 6A18      M    MOVFP    DPX+B0,WREG          ; get lowest byte of a into w
007E 0F44      M    ADDWF    ACC+B0, F          ; add lowest byte of b, save in b(B0)
007F 6A19      M    MOVFP    DPX+B1,WREG          ; get 2nd byte of a into w
0080 1145      M    ADDWFC    ACC+B1, F          ; add 2nd byte of b, save in b(B1)
0081 6A1A      M    MOVFP    DPX+B2,WREG          ; get 3rd byte of a into w
0082 1146      M    ADDWFC    ACC+B2, F          ; add 3rd byte of b, save in b(B2)
0083 6A1B      M    MOVFP    DPX+B3,WREG          ; get 4th byte of a into w
0084 1147      M    ADDWFC    ACC+B3, F          ; add 4th byte of b, save in b(B3)
M
M ;
M ;
M ; save the upper 16 bits of D(n) from the 32 bit accumulator
M ; left shift the result by 1, to adjust the decimal point after
M ; a Q15*Q15 multiplication
M ;
0085 1A45      M    rlcfc    ACC+B1,w
0086 1A46      M    rlcfc    ACC+B2,w
0087 0126      M    movwf    Dn2
0088 1A47      M    rlcfc    ACC+B3,w          ; decimal adjust ( mult by 2)
0089 0127      M    movwf    Dn2+B1
M ;
M ; Compute B2 * [D(n) + D(n-2)]
M ;
M    #if B0_EQUALS_B2
M
M    ADD16ACC   Dn2_2,Dn2,AARG          ; AARG = Dn + D(n-2) = multiplier
M
008A 6A2A      M    movfp    Dn2_2+B0,WREG
008B 0E26      M    addwf    Dn2+B0,w
008C 011C      M    movwf    AARG+B0
008D 6A2B      M    movfp    Dn2_2+B1,WREG
008E 1027      M    addwfc    Dn2+B1,w
008F 011D      M    movwf    AARG+B1
M
M    MOVFP16   B20,BARG          ; BARG = A2 = multiplicand
M

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0090 7E3A      M   MOVFP   B20+B0,BARG+B0      ; move A(B0) to B(B0)
0091 7F3B      M   MOVFP   B20+B1,BARG+B1      ; move A(B1) to B(B1)
M
0092 E0AF      M   call    Db1Mult              ; (ACCd,ACCe) = B2*[D(n)+D(n-2)]
M   MOVFP32   DPX,ACC
M
0093 5844      M   MOVFP   DPX+B0,ACC+B0        ; move A(B0) to B(B0)
0094 5945      M   MOVFP   DPX+B1,ACC+B1        ; move A(B1) to B(B1)
0095 5A46      M   MOVFP   DPX+B2,ACC+B2        ; move A(B2) to B(B2)
0096 5B47      M   MOVFP   DPX+B3,ACC+B3        ; move A(B3) to B(B3)
M
M
M   #else
M
M   MOVFP16   B20,BARG
M   MOVFP16   Dn2,AARG
M   call      Db1Mult              ; B0*D(n)
M   MOVFP32   DPX,ACC
M
M   MOVFP16   Bx2,BARG
M   MOVFP16   Dn2_2,AARG
M   call      Db1Mult              ; B2*D(n-2)
M   ADD32     DPX,ACC
M
M   #endif
M ;
M ; Shift down D(n-1) to D(n-2) after D(n-2) usage is no longer required.
M ; This way in the next iteration D(n-2) is equal to the present D(n-1)
M ;
0097 7C28      M   movfp   Dn2_1,AARG+B0
0098 5C2A      M   movpf   AARG+B0,Dn2_2        ; Shift down D(n-1)
0099 7D29      M   movfp   Dn2_1+B1,AARG+B1
009A 5D2B      M   movpf   AARG+B1,Dn2_2+B1      ; AARG = D(n-1) = multiplier
M
M   MOVFP16   B21,BARG              ; BARG = B1 = multiplicand
M
009B 7E3C      M   MOVFP   B21+B0,BARG+B0        ; move A(B0) to B(B0)
009C 7F3D      M   MOVFP   B21+B1,BARG+B1        ; move A(B1) to B(B1)
M
M
009D E0AF      M   call      Db1Mult              ; (ACCd,ACCe) = B1*D(n-1)
M ;
M ; Compute Output Y = B1*D(n-1) + B2*D(n-2) + B0*D(n)
M ;                      = B1*D(n-1) + B0*[D(n) + D(n-2)]
M ; Since all multiplications are already done, simply perform a
M ; 32 bit addition
M ;
M   ADD32     DPX,ACC              ; ACC = B1*D(n-1) + B2*D(n-2) + B0*D(n)
M
009E 6A18      M   MOVFP   DPX+B0,WREG          ; get lowest byte of a into w
009F 0F44      M   ADDWF   ACC+B0, F            ; add lowest byte of b, save in b(B0)
00A0 6A19      M   MOVFP   DPX+B1,WREG          ; get 2nd byte of a into w
00A1 1145      M   ADDWFC  ACC+B1, F            ; add 2nd byte of b, save in b(B1)
00A2 6A1A      M   MOVFP   DPX+B2,WREG          ; get 3rd byte of a into w
00A3 1146      M   ADDWFC  ACC+B2, F            ; add 3rd byte of b, save in b(B2)
00A4 6A1B      M   MOVFP   DPX+B3,WREG          ; get 4th byte of a into w
00A5 1147      M   ADDWFC  ACC+B3, F            ; add 4th byte of b, save in b(B3)
M
M ;
M ; Shift down D(n) to D(n-1) so that in the next iteration, the new
M ; D(n-1) is the present D(n)
M ;
M   MOV16     Dn2,Dn2_1            ; Shift down D(n) to D(n-1)
M
00A6 6A26      M   MOVFP   Dn2+B0,WREG          ; get byte of a into w
00A7 0128      M   MOVWF   Dn2_1+B0            ; move to b(B0)

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00A8 6A27      M  MOVFP   Dn2+B1,WREG          ; get byte of a into w
00A9 0129      M  MOVWF   Dn2_1+B1           ; move to b(B1)
               M
               M ;
00269 ;
00270 ; The filter output is now computed
00271 ; Save the Upper 16 Bits of 32 bit Accumulator into Y after
00272 ; adjusting the decimal point
00273 ;
00274 ;
00275 MOV16     ACC+BB2,Y
               M
00AA 6A46      M  MOVFP   ACC+BB2+B0,WREG      ; get byte of a into w
00AB 0142      M  MOVWF   Y+B0                ; move to b(B0)
00AC 6A47      M  MOVFP   ACC+BB2+B1,WREG      ; get byte of a into w
00AD 0143      M  MOVWF   Y+B1                ; move to b(B1)
               M
00276 ;
00AE 0002      00277  return                    ; Output Y(n) computed
00278 ;
00279 ;*****
00280 ; Set SIGNED/UNSIGNED Flag Before Including 17c42MPY.mac
00281 ;
00282 #include    <17c42MPY.mac>
00178 LIST
00283
00284 ;*****
00285 ; Low Pass Filter Co-efficients
00286 ;
00287 ;
00288 ; ELLIPTIC      LOWPASS FILTER
00289 ;
00290 ; FILTER ORDER = 4
00291 ; Sampling frequency = 2.000 KiloHertz
00292 ;
00293 ; BAND 1      BAND 2
00294 ;
00295 ;
00296 ; LOWER BAND EDGE      .00000      .60000
00297 ; UPPER BAND EDGE    .50000      1.00000
00298 ; NOMINAL GAIN        1.00000      .00000
00299 ; NOMINAL RIPPLE      .01000      .05000
00300 ; MAXIMUM RIPPLE     .00906      .04601
00301 ; RIPPLE IN dB       .07830      -26.74235
00302 ;
00303 ; I      A(I,1)      A(I,2)      B(I,0)      B(I,1)      B(I,2)
00304 ;
00305 ; 1      -.133331     .167145     .285431     .462921     .285431
00306 ;      .147827      .765900     .698273     .499908     .698273
00307 ;
00308
01C1      00309 _coeff_lpass          ; co-efficients for 1st Cascade section
01C1 1111    00310 data      4369          ; -A11
01C2 EA9B    00311 data     -5477          ; -A12
01C3 2489    00312 data      9353          ; B10
01C4 3B41    00313 data     15169          ; B11
01C5 2489    00314 data      9353          ; B12
00315          ; co-efficients for 2nd Cascade section
01C6 ED14    00316 data     -4844          ; -A21
01C7 9DF7    00317 data    -25097          ; -A22
01C8 5961    00318 data     22881          ; B20
01C9 3FFD    00319 data     16381          ; B21
01CA 5961    00320 data     22881          ; B22
00321 ;
00322 ;*****
00323 ;

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00324 ;
00325 ;*****
00326 ; High Pass Filter Co-efficients
00327 ;
00328 ;
00329 ;          ELLIPTIC    HIGHPASS FILTER
00330 ;
00331 ;          FILTER ORDER =    4
00332 ;          Sampling frequency =    2.000 KiloHertz
00333 ;
00334 ;          BAND 1          BAND 2
00335 ;
00336 ;
00337 ; LOWER BAND EDGE          .00000          .50000
00338 ; UPPER BAND EDGE          .40000          1.00000
00339 ; NOMINAL GAIN          .00000          1.00000
00340 ; NOMINAL RIPPLE          .04000          .02000
00341 ; MAXIMUM RIPPLE          .03368          .01686
00342 ; RIPPLE IN dB          -29.45335          .14526
00343 ;
00344 ; I          A(I,1)          A(I,2)          B(I,0)          B(I,1)          B(I,2)
00345 ;
00346 ; 1          .276886          .195648          .253677          -.411407          .253677
00347 ; 2          -.094299          .780396          .678650          -.485840          .678650
00348 ;
00349 ;
01CB 00350 _coeff_hpass          ; co-efficients for 1st Cascade section
01CB DC8F 00351 data          -9073          ; -A11
01CC E6F5 00352 data          -6411          ; -A12
01CD 2079 00353 data          8313          ; B10
01CE CB57 00354 data          -13481          ; B11
01CF 2079 00355 data          8313          ; B12
00356 ;          ; co-efficients for 2nd Cascade section
01D0 0C12 00357 data          3090          ; -A21
01D1 9C1C 00358 data          -25572          ; -A22
01D2 56DE 00359 data          22238          ; B20
01D3 C1D0 00360 data          -15920          ; B21
01D4 56DE 00361 data          22238          ; B22
00362 ;
00363 ;*****
00364 ;
00365 ;*****
00366 ; Band Pass Filter Co-efficients
00367 ;
00368 ;
00369 ;          ELLIPTIC    BANDPASS FILTER
00370 ;
00371 ;          FILTER ORDER =    4
00372 ;          Sampling frequency =    2.000 KiloHertz
00373 ;
00374 ;          BAND 1          BAND 2          BAND 3
00375 ;
00376 ;
00377 ; LOWER BAND EDGE          .00000          .30000          .90000
00378 ; UPPER BAND EDGE          .10000          .70000          1.00000
00379 ; NOMINAL GAIN          .00000          1.00000          .00000
00380 ; NOMINAL RIPPLE          .05000          .05000          .05000
00381 ; MAXIMUM RIPPLE          .03644          .03867          .03641
00382 ; RIPPLE IN dB          -28.76779          .32956          -28.77647
00383 ;
00384 ;
00385 ; I          A(I,1)          A(I,2)          B(I,0)          B(I,1)          B(I,2)
00386 ;
00387 ; 1          -.936676          .550568          .444000          -.865173          .444000
00388 ; 2          .936707          .550568          .615540          1.199402          .615540
00389 ;

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00390
01D5      00391 _coeff_bpasp           ; co-efficients for 1st Cascade section
01D5 3BF2 00392      data      30693/2      ; -A11
01D6 DCC4 00393      data      -18041/2     ; -A12
01D7 1C6A 00394      data      14549/2     ; B10
01D8 C8A1 00395      data      -28350/2    ; B11
01D9 1C6A 00396      data      14549/2     ; B12
00397 ;
                                ; co-efficients for 2nd Cascade section
01DA C40D 00398      data      -30694/2     ; -A21
01DB DCC4 00399      data      -18041/2     ; -A22
01DC 2765 00400      data      20170/2     ; B20
01DD 4CC3 00401      data      39302/2     ; B21
01DE 2765 00402      data      20170/2     ; B22
00403 ;
00404 ;*****
00405 ;
00406 ;*****
00407 ; Band Stop Filter Co-efficients
00408 ;
00409 ;
00410 ; ELLIPTIC BANDSTOP FILTER
00411 ;
00412 ; FILTER ORDER = 4
00413 ; Sampling frequency = 2.000 KiloHertz
00414 ;
00415 ; BAND 1 BAND 2 BAND 3
00416 ;
00417 ;
00418 ; LOWER BAND EDGE .00000 .45000 .70000
00419 ; UPPER BAND EDGE .30000 .55000 1.00000
00420 ; NOMINAL GAIN 1.00000 .00000 1.00000
00421 ; NOMINAL RIPPLE .05000 .05000 .05000
00422 ; MAXIMUM RIPPLE .03516 .03241 .03517
00423 ; RIPPLE IN dB .30015 -29.78523 .30027
00424 ;
00425 ;
00426 ; I A(I,1) A(I,2) B(I,0) B(I,1) B(I,2)
00427 ;
00428 ; 1 .749420 .583282 .392685 .087936 .392685
00429 ; 2 -.749390 .583282 1.210022 -.270935 1.210022
00430 ;
01DF      00431 _coeff_bstop           ; co-efficients for 1st Cascade section
01DF D00A 00432      data      -24557/2     ; -A11
01E0 DAAC 00433      data      -19113/2     ; -A12
01E1 1922 00434      data      12868/2     ; B10
01E2 05A0 00435      data      2881/2      ; B11
01E3 1922 00436      data      12868/2     ; B12
00437 ;
                                ; co-efficients for 2nd Cascade section
01E4 2FF6 00438      data      24556/2     ; -A21
01E5 DAAC 00439      data      -19113/2     ; -A22
01E6 4D71 00440      data      39650/2     ; B20
01E7 EEA9 00441      data      -8878/2     ; B21
01E8 4D71 00442      data      39650/2     ; B22
00443 ;
00444 ;*****
00445 ;
00446 END

```

MEMORY USAGE MAP ('X' = Used, '-' = Unused)

```

0000 : XXXXXXXXXXXXXXXXXXXX XXXXXXXXXXXXXXXXXXXX XXXXXXXXXXXXXXXXXXXX XXXXXXXXXXXXXXXXXXXX
0040 : XXXXXXXXXXXXXXXXXXXX XXXXXXXXXXXXXXXXXXXX XXXXXXXXXXXXXXXXXXXX XXXXXXXXXXXXXXXXXXXX
0080 : XXXXXXXXXXXXXXXXXXXX XXXXXXXXXXXXXXXXXXXX XXXXXXXXXXXXXXXXXXXX XXXXXXXXXXXXXXXXXXXX
00C0 : XXXXXXXXXXXXXXXXXXXX XXXXXXXXXXXXXXXXXXXX XXXXXXXXXXXXXXXXXXXX XXXXXXXXXXXXXXXXXXXX
0100 : XXXXXXXXXXXXXXXXXXXX XXXXXXXXXXXXXXXXXXXX XXXXXXXXXXXXXXXXXXXX XXXXXXXXXXXXXXXXXXXX
0140 : XXXXXXXXXXXXXXXXXXXX XXXXXXXXXXXXXXXXXXXX XXXXXXXXXXXXXXXXXXXX XXXXXXXXXXXXXXXXXXXX
0180 : XXXXXXXXXXXXXXXXXXXX XXXXXXXXXXXXXXXXXXXX XXXXXXXXXXXXXXXXXXXX XXXXXXXXXXXXXXXXXXXX
01C0 : XXXXXXXXXXXXXXXXXXXX XXXXXXXXXXXXXXXXXXXX XXXXXXXXXX-----

```

All other memory blocks unused.

Program Memory Words Used: 489

```

Errors      : 0
Warnings    : 0 reported, 0 suppressed
Messages    : 0 reported, 0 suppressed

```

---

Note the following details of the code protection feature on PICmicro® MCUs.

- The PICmicro family meets the specifications contained in the Microchip Data Sheet.
- Microchip believes that its family of PICmicro microcontrollers is one of the most secure products of its kind on the market today, when used in the intended manner and under normal conditions.
- There are dishonest and possibly illegal methods used to breach the code protection feature. All of these methods, to our knowledge, require using the PICmicro microcontroller in a manner outside the operating specifications contained in the data sheet. The person doing so may be engaged in theft of intellectual property.
- Microchip is willing to work with the customer who is concerned about the integrity of their code.
- Neither Microchip nor any other semiconductor manufacturer can guarantee the security of their code. Code protection does not mean that we are guaranteeing the product as "unbreakable".
- Code protection is constantly evolving. We at Microchip are committed to continuously improving the code protection features of our product.

If you have any further questions about this matter, please contact the local sales office nearest to you.

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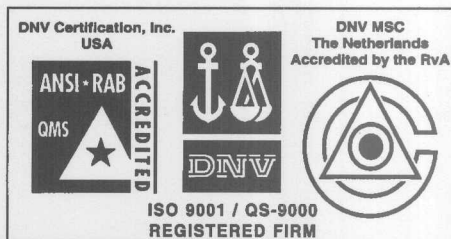
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